

Alcatel 4400 Rel3.2 PBX with Cisco CallManager (CCM) Using 6608-E1 QSIG Protocol as MGCP Gateway

This application note illustrates connectivity for Alcatel 4400 Release 3.2 PBX with Cisco CallManager using Cisco 6608-E1 QSIG as MGCP Gateway.

Introduction

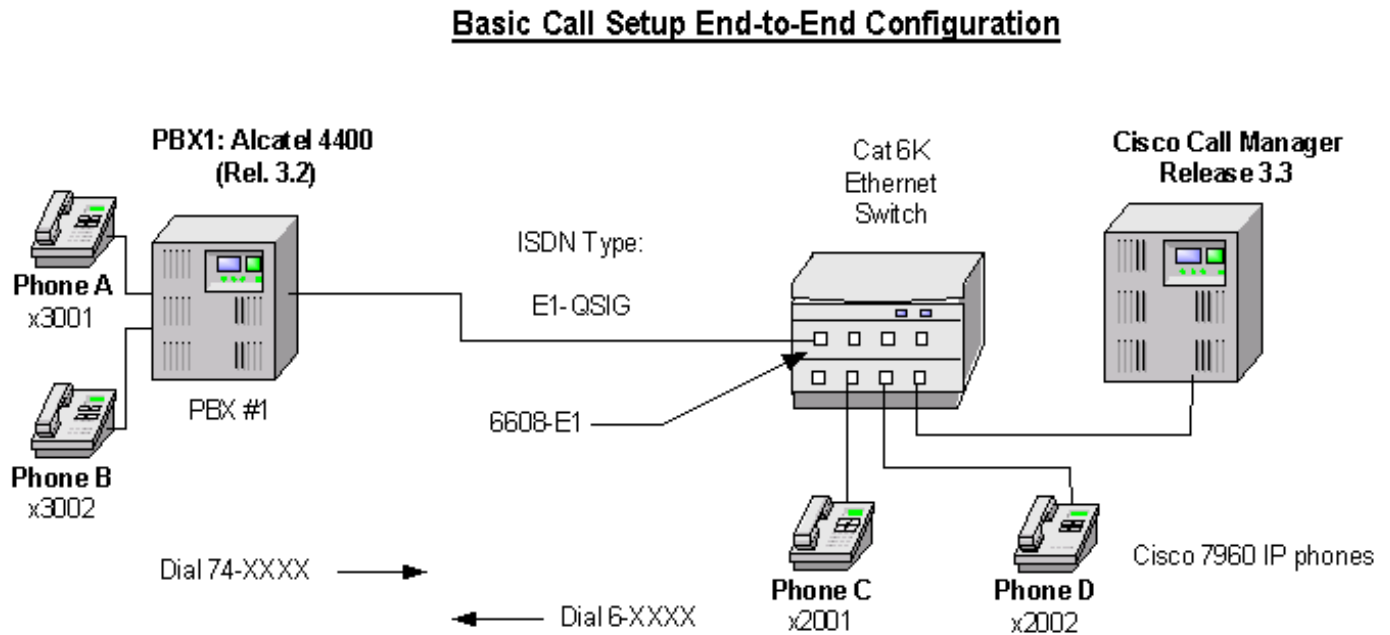
Key attributes of the network environment used for this application note:

- The network topology diagram shows the test set-up for end-to-end interoperability with the Cisco CallManager (CCM) connected to the PBX via 6608-E1 QSIG link as MGCP Gateway.
- Connectivity is achieved by using the PRI ISO QSIG E1 protocol type on the MGCP gateway and ABC-F (ISO version) as protocol type on the Alcatel 4400 PBX.
- Basic Calls worked fine in both directions with Calling/Connected Name and Calling Number features support.
- Alcatel 4400 PBX does not honor Connected Number Identification Presentation feature. Alcatel phone does not display *Connected Number* even though CCM is sending the information in the connect message.
- Call Forwarding did not work when the IP phone calls an Alcatel phone that is setup to forward to another Alcatel phone.



Network Topology

Figure 1
Network Topology or Test Setup



Limitations

Calling Name and Number Feature

When calling from Alcatel digital phone to Cisco 7960 IP phone, the Cisco IP phone displays Calling Name and Number when the call is answered. Alcatel phone however displays connected Name only, No connected Number when the call is answered. Alcatel does not honor Connected Number Identification Presentation QSIG service even though CCM is sending the Connected Number information in the connect message.

CCM Seaview release does not support sending Alerting Name or Busy Name Identification information.

Call Forwarding

Calls did not complete when the IP phone calls an Alcatel phone that is setup to forward to another Alcatel phone. Alcatel in this case sends Facility IE for 'CallRerouting'. Since CCM3.3 Seaview release does not support CallRerouting, the call is not completed.

System Components

Hardware Requirements

Cisco Hardware:



- Cisco Cat6K switch with 6608-E1 Gateway
- Cisco CM 3.3

Alcatel 4400 PBX Hardware:

- PRA2, 3BA23076- E1 PRI/QSIG

Software Requirements

Alcatel: PBX Software version 3.2

Cisco: Cisco CM 3.3

Features

Key features supported:

- Calling Name Identification Presentation
- Calling Number Identification Presentation
- Calling Number Identification Restriction
- Connected Name Identification Presentation
- Connected Number Identification Presentation
- Honor Calling/Connected Name Identification Restriction
- Honor Calling/Connected Number Identification Restriction
- Key features not supported:
- Sending Alerting Name Identification
- Sending Busy Name Identification
- Sending Calling/Connected Name Identification Restriction
- Sending Connected Number Identification Restriction
- Updating Connected Number on Alcatel phone for Basic Calls
- Updating Connected Name and Number for Call Transfers



- Updating Connected Name and Number for Call Forwarding

Configuring the Alcatel 4400 PBX

Configure in the following sequence:

1. [“Configure “ISO Function” System Parameters” on page 4](#)
2. [“Configure Board” on page 5](#)
3. [“Configure Digital Access Options” on page 10](#)
4. [“Configure Trunk Group” on page 12](#)
5. [“Configure Trunk Detail” on page 13](#)
6. [“Configure T2 Access” on page 15](#)

Configure “ISO Function” System Parameters

Command: `\compidea\System::1\Other System Param::1`

Table 1 Other System Parameter Listing

| Parameter | Value |
|-----------------------------------|-------|
| Instance (reserved) | 1 |
| Trunk seizure via attendant | True |
| No detect.of On-hook tone | True |
| TrkGrp in ticket for trans.call | True |
| VPN service | False |
| ISVPN Node No. | 1 |
| Nb Digits displayed on sets | 1 |
| Melody Ringing Type | 1 |
| Int.Call Ringing Cadence No | 1 |
| Ext.Call Ringing Cadence No | 1 |
| Executive Type Ringing Cadence No | 1 |
| Prioritary Call Cadence No | 1 |
| ISO Function | True |
| Booking B Channel | True |
| Uncontrol.Business Account Code | False |
| Business Pref.With Business No. | True |
| Project prefix With Code | False |
| Follow-Me on Remote forwarding | False |
| BC HLC Fax | 1 |



Table 1 Other System Parameter Listing

| Parameter | Value |
|-----------------------------------|-------|
| VG Recording Gain from an UA set | 3 |
| Calling Id.length | 7 |
| Nb Of Secret Code Errors | 0 |
| Transfer All Business Call Types | True |
| Attendt Keep Local on Enqu.cancel | False |
| Compatibility GF | True |
| Alphanum.Char.Entering-mode2 | True |
| Spain version 2 | False |
| Delete First STAR Digit | True |
| QSIG1 (reserved) | 0 |
| QSIG2 (reserved) | 0 |
| Stop Tie Line Supervision | False |
| Tie Line Germany | 0 |
| Entity For Virtual Set | 0 |
| Nb Of Business Code Errors | 0 |
| Disabled Code Duration | 0 |
| Poor ARS Rerouting memo.Dur. | 20 |
| Poor ARS Route Inhibit Dur. | 180 |
| Charging by Nb MiniMessages | True |
| Remote Numeric Gain For 4630 | True |
| Project Code In Redial Key | True |
| Cx on Progress message | False |
| Differred transm.(Swiss trick) | False |
| Send NDS NDI | False |
| Calls Distributed On Att.In Order | False |
| Nb non answ. int. messages by set | 16 |
| NS_read_before_ack | False |
| SNCM | 0 |

Configure Board

Note: Interface type must be set to PRA2



Command: \compidea\Shelf::0\Board::4

Table 2 Board Parameter Listings

| Board Parameter | Value |
|-------------------------------|---------------|
| Board Address | 4 |
| Interface Type | PRA2 |
| Administrative status | Enabled |
| Usage State | Busy |
| Operational State | Enabled |
| Main/Standby State | Main (Master) |
| Number Of Sets Being Connect. | 1 |
| Remote Shelf Address | 255 |
| Remote Board Address | 255 |
| Synchronisation Priority | 255 |
| IO2 With SPB | NO |
| AUXU Parameters 1 | None |
| AUXU Parameters 2 | None |
| AUXU Parameters 3 | None |
| AUXU Parameters 4 | None |
| CRC4 | YES |
| Country Protocol Type | Default |
| Time Slots\0 | 0 |
| Time Slots\1 | 1 |
| Time Slots\2 | 1 |
| Time Slots\3 | 1 |
| Time Slots\4 | 1 |
| Time Slots\5 | 1 |
| Time Slots\6 | 1 |
| Time Slots\7 | 1 |
| Time Slots\8 | 1 |



Table 2 Board Parameter Listings

| Board Parameter | Value |
|---------------------------|-------------|
| Time Slots\9 | 1 |
| Time Slots\10 | 1 |
| Time Slots\11 | 1 |
| Time Slots\12 | 1 |
| Time Slots\13 | 1 |
| Time Slots\14 | 1 |
| Time Slots\15 | 1 |
| Time Slots\16 | 0 |
| Time Slots\17 | 1 |
| Time Slots\18 | 1 |
| Time Slots\19 | 1 |
| Time Slots\20 | 1 |
| Time Slots\21 | 1 |
| Time Slots\22 | 1 |
| Time Slots\23 | 1 |
| Time Slots\24 | 1 |
| Time Slots\25 | 1 |
| Time Slots\26 | 1 |
| Time Slots\27 | 1 |
| Time Slots\28 | 1 |
| Time Slots\29 | 1 |
| Time Slots\30 | 1 |
| Time Slots\31 | 1 |
| Voice-->Data TS | YES |
| SU shelf Type | 2 PCM Shelf |
| DECT Location area number | 255 |
| Send Init Dynamic Msg | False |
| Param By Default | True |
| Clock Mode | Internal |



Table 2 Board Parameter Listings

| Board Parameter | Value |
|---|-------|
| CPU with Optimized B ChannelAccess | NO |
| Board with DTM | False |
| Incidents Teleservice | YES |
| Max.VG Recording Duration | 0 |
| DASS2 Simulate Network | NO |
| DPNSS Layer 2 Address | A |
| ISDN Board Layer 2 Parameters\Retransmission Timer | 100 |
| ISDN Board Layer 2 Parameters\TEI Identity Check Timer | 100 |
| ISDN Board Layer 2 Parameters\Polling Timer | 1000 |
| ISDN Board Layer 2 Parameters\Nb_Of_Retransmission | 3 |
| ISDN Board Layer 2 Parameters\Max Frame Size (Bytes) | 260 |
| ISDN Board Layer 2 Parameters\Window Size In Frames SAPI S T0 | 1 |
| ISDN Board Layer 2 Parameters\Window Size In Frames SAPI P T0 | 3 |
| ISDN Board Layer 2 Parameters\Window Size In Frames SAPI S T2 | 7 |
| ISDN Board Layer 2 Parameters\Window Size In Frames SAPI P T2 | 7 |
| Number of configured ports | 1 |



Table 2 Board Parameter Listings

| Board Parameter | Value |
|-------------------------------|-----------------|
| Associated CPU | 255 |
| Number of configured E1 ports | 8 |
| Synchronisation mode | Adaptive method |
| In Band Signalling | NO |
| Passive board | NO |
| SS7 signalling | NO |
| PRA7 TS signalling | 16 |
| Use Data Compression | NO |
| Mutual Aid | YES |
| LIO Daughter Board | 6 Compressors |
| Tone on Board | R2 Tone |
| Number of Used Compressors | 0 |
| GNISC in Rack | 255 |
| GNISC in position | 255 |
| Usage State | Suite Slave |
| Atm address | |
| TS used on PCM 0 | 0 |
| TS used on PCM 1 | 0 |
| TS used on PCM 2 | 0 |
| TS used on PCM 3 | 0 |
| TS used on PCM 4 | 0 |
| TS used on PCM 5 | 0 |
| TS used on PCM 6 | 0 |
| TS used on PCM 7 | 0 |
| Daughter board equipped | NO |
| Number of Used Compressors | 0 |
| Mode | Gateway IP |



Table 2 Board Parameter Listings

| Board Parameter | Value |
|-------------------------|--------------------------|
| Embedded Ethernet | YES |
| Voice Guide Lang Index | 1 |
| CLIP Signalization | No CLIP |
| IVR Protocol | No IVR Protocol Protocol |
| 4615 Present | NO |
| LIOE coupler 1 address | 255 |
| LIOE coupler 2 address | 255 |
| Associated BBC2 coupler | 255 |
| Associated BBC2 access | 255 |
| Use of volume in system | YES |
| Local volume (dB) | 0 |

Configure Digital Access Options

Note: Network mode must be set to Yes for (Master/Network) or No- (Slave/User); Access Type must be set to T2.

Command: `\compidea\Shelf::0\Board::4\Digital Access::0`

Table 3 Digital Access Options Listing

| Parameter | Value |
|---------------------------------|-------|
| T0/T2 Access No. | 0 |
| Access Type | T2 |
| Used Access | YES |
| Synchronisation Priority | 255 |
| Network Mode | YES |
| Max Nb Of Used B Channels | 30 |
| Max_Nb_Of_Compressed_B_Channels | 0 |



Table 3 Digital Access Options Listing

| Parameter | Value |
|------------------------------|---------------------------|
| Nb Of Signalization TS | 1 |
| TieLine Mode | YES |
| With Alarm | NO |
| Access Type S0 | NO |
| Reserved1 | NO |
| Reserved2 | NO |
| Network Date Time Update | NO |
| CRC4 | YES |
| Port Class | NOT SIG |
| Multiframe Type | SF |
| Line Type | Short Haul 0 to 35 meters |
| Pulses Encoding | AMI |
| Retransmission Timer | 100 |
| TEI Identity Check Timer | 100 |
| Polling Timer | 1000 |
| Nb_Of_Retransmission | 3 |
| Max Frame Size (Bytes) | 260 |
| Window Size In Frames SAPI S | 7 |
| Window Size In SAPI P | 7 |
| B Channel Rate | 64K |
| | 64K |



Configure Trunk Group

Note: Q931 signal variant is used to set the protocol type to ABC-F

Command: \compidea\Trunk Groups::1

Table 4 Trunk Group Parameter Listing

| Parameter | Value |
|--------------------------------|----------|
| Trunk Groups | |
| Trunk Group Id | 1 |
| Trunk Group Type | T2 |
| Trunk Group Name | PRI-ABCF |
| Node number | 1 |
| Transcom Trunk Group | False |
| Auto.reserv.by Attendant | False |
| Overflow trunk group No. | -1 |
| Tone on seizure | False |
| Private Trunk Group | False |
| Paging Trunk Group | False |
| Paging Table Id | -1 |
| Paging Signalization | NDDI |
| Security Patrol | False |
| Q931 signal variant | ABC-F |
| Operator Id | ANSI |
| Number Compatible With | -1 |
| Prefix Sending | False |
| Number Of Digits To Send | 4 |
| Channel selection type | Quantum |
| Remote Network | 15 |
| Shared Trunk Group | False |
| T.line Calling last dig.length | 0 |



Table 4 Trunk Group Parameter Listing

| Parameter | Value |
|------------------------------------|----------------|
| auto.DTMF dialing on outgoing call | NO |
| T2 Specificity | None |
| Public Network Category | 0 |
| DDI transcoding | False |
| Special Services | Nothing |
| Can support UUS in SETUP | True |
| Register Signalling | Decadic/MF Q23 |

Configure Trunk Detail

Command: \compidea\Trunk Groups::1\Trunk Group::1

Table 5 Trunk Detail Listing

| Parameter | Value |
|----------------------------|-------|
| Trunk Group | |
| Instance (reserved) | 1 |
| Trunk Group Type | T2 |
| Public Network Ref. | |
| Dialling end to end | NO |
| DTMF end to end signal. | NO |
| Paying Incoming Calls | NO |
| TS Permanently assigned | NO |
| Min. Nb.of digits on seize | 0 |
| Signal.with access code | NO |
| Trunk group used in DISA | NO |
| DISA Secret Code | |



Table 5 Trunk Detail Listing

| Parameter | Value |
|------------------------------------|-------|
| VG for non-existent No. | YES |
| Routing To Executive | NO |
| Trunk Category Id | 19 |
| Nb of digits unused (ISDN) | 4 |
| B Channel Choice | NO |
| Channels Reserved By Attend. | 0 |
| Dissuasion For ACD | NO |
| DTO joining | NO |
| Enquiry Call On B Channel | NO |
| DDI Mode | NO |
| Automated Attendant | NO |
| Calling party Rights category | 0 |
| Entity Number | 0 |
| TS Overflow | YES |
| Number To Be Added | |
| Supervised by Routing | NO |
| Access Cluster Id | -1 |
| VPN Cost Limit for Incom.Calls | 0 |
| Immediat Trk Listening For VPNCall | YES |
| VPN TS % | 50 |
| Csta Monitored | NO |
| Max.% of trunks out CCD | 0 |
| Charge Calling And ADN Creation | NO |



Table 5 Trunk Detail Listing

| Parameter | Value |
|--|----------------|
| Ratio analog.to ISDN tax | |
| Collect Calls Allowed | YES |
| Priority of Call | NO |
| PCM Network Mode | NO |
| LogicalChannel | 1__15 & 17__31 |
| TS Distribution on Accesses | YES |
| Use Split Acces | NO |
| Heterogeneous Remote Network | NO |
| Barring mode | Not barred |
| ARS class of service | 31 |
| Quality profile for voice on IP | Profile #1 |
| IP compression type | Default |
| Use of volume in system | YES |
| Local volume (dB) | 0 |

Configure T2 Access

Command: \compidea\Trunk Groups::1\T2/T1/T0 Access::0-4-0

Table 6 T2 Access Parameter Listing

| Parameter | Value |
|---------------------------|----------------------------------|
| T2/T1/T0 Access | |
| Physical Address | 0-4-0 |
| Access Type | T2 |
| Access Cluster Id | -1 |
| Time Slots T2 | 01111111111111110111111111111111 |
| Time Slots T0 | 011 |
| Time Slots T1 CCS | 01111111111111111111111111111110 |
| Time Slots Virtual | 01111110000000 |



Table 6 T2 Access Parameter Listing

| Parameter | Value |
|-------------------------------|-------|
| DLCI | 16 |
| Committed Information Rate | 48 |
| Extended Information Rate | 64 |
| CIR Measurement Interval | 10 |
| Support Time Slot Address | |
| ISDN Compression Number | |
| Release Support Timer (100ms) | 0 |
| Protection Timer (1mn) | 0 |



Configuring CCM—6608-E1 Gateway Configuration

Cisco CallManager Administration
For Cisco IP Telephony Solutions

Gateway Configuration [Back to Find/List Gateways](#)

Product : Cisco Catalyst 6000 E1 VoIP Gateway
Gateway : S0/DS1-0@SDA000164122280
Device Protocol: Digital Access PRI
Registration: Registered with Cisco CallManager 10.10.10.1
IP Address: [10.10.10.104](#)

Status: Ready

Device Information

| | |
|---------------------------|--|
| MAC Address* | <input type="text" value="000164122280"/> |
| Description | <input type="text" value="SDA000164122280"/> |
| Device Pool* | <input type="text" value="Default"/> |
| Network Locale* | <input type="text" value="United States"/> |
| Media Resource Group List | <input type="text" value="< None >"/> |
| Location | <input type="text" value="< None >"/> |
| AAR Group | <input type="text" value="< None >"/> |
| Load Information | <input type="text"/> |

Interface Information

| | |
|--------------------------|--|
| PRI Protocol Type* | <input type="text" value="PRI ISO QSIG E1"/> |
| Protocol Side* | <input type="text" value="User"/> |
| Channel Selection Order* | <input type="text" value="Top Down"/> |



Cisco CallManager 3.3 Administration - Gateway Configuration - Microsoft Internet Explorer

File Edit View Favorites Tools Help

Back Forward Stop Home Search Favorites History Print

Address <http://ferengi/CCMAdmin/gatewayconfig.asp?plid={CB557D64-12B9-451A-979A-E4D32198D86A}&type=2> Go Links

| | |
|--|-----------------|
| Channel IE Type* | Timeslot Number |
| PCM Type* | A-law |
| Delay for first restart (1/8 sec ticks) | 32 |
| Delay between restarts (1/8 sec ticks) | 4 |
| <input checked="" type="checkbox"/> Inhibit restarts at PRI initialization | |
| <input type="checkbox"/> Enable status poll | |

Call Routing Information

Inbound Calls

| | |
|--------------------------|----------|
| Significant Digits* | 23 |
| Calling Search Space | < None > |
| AAR Calling Search Space | < None > |
| Prefix DN | |

Outbound Calls

| | |
|---------------------------------------|-------------------|
| Calling Party Presentation* | Allowed |
| Calling Party Selection* | Originator |
| Called party IE number type unknown* | Cisco CallManager |
| Calling party IE number type unknown* | Cisco CallManager |
| Called Numbering Plan* | Private |
| Calling Numbering Plan* | Cisco CallManager |
| Number of digits to strip* | 0 |
| Caller ID DN | |

PRI Protocol Type Specific Information

| |
|--|
| <input type="checkbox"/> Display IE Delivery |
|--|

Local intranet



Cisco CallManager 3.3 Administration - Gateway Configuration - Microsoft Internet Explorer

File Edit View Favorites Tools Help

Back Forward Stop Home Search Favorites History Print

Address <http://ferengi/CCMAdmin/gatewayconfig.asp?pkid={CB557D64-12B9-451A-979A-E4D32198D86A}&type=2> Go Links

Redirecting Number IE Delivery - Outbound

Redirecting Number IE Delivery - Inbound

Send Extra Leading Character In DisplayIE***

Setup non-ISDN Progress Indicator IE Enable****

MCDN Channel Number Extension Bit Set to Zero**

Interface Identifier Present**

Interface Identifier Value**

Product Specific Configuration

Clock Reference*

Framing*

Audio Signal Adjustment into IP Network*

Audio Signal Adjustment from IP Network*

Zero Suppression*

Digit On Duration(50-500ms)*

Interdigit Duration(50-500msec)*

Adaptive Gain Control Enable*

SNMP Community String

Fax Parameters

Fax Relay Enable*

Fax Error Correction Mode Override*

* indicates required item

** applicable to DMS-100 protocol only

*** applicable to DMS-100 protocol and DMS-250 protocol only

Local Intranet



Enbloc Route Pattern Configuration

Cisco CallManager 3.3 Administration - Route Pattern Configuration - Microsoft Internet Explorer

File Edit View Favorites Tools Help

Back Forward Stop Home Search Favorites History Print

Address <http://ferengi/CCMAdmin/routepatternconfig.asp?pkid={AB9DD1D1-A861-44F8-A286-72E0CA4426A1}> Go Links

Route Pattern: 6.XXXX

Status: Ready
Note: Any update to this route pattern automatically resets the associated gateway/route list

Copy Update Delete

Pattern Definition

Route Pattern*

Partition

Description

Numbering Plan*

Route Filter

Gateway/Route List* (Edit)

Route Option Route this pattern Block this pattern

Provide Outside Dial Tone Urgent Priority

Calling Party Transformations

Use Calling Party's External Phone Number Mask

Calling Party Transform Mask

Prefix Digits (Outgoing Calls)

Calling Party Presentation

Called Party Transformations

Discard Digits

Called Party Transform Mask

Prefix Digits (Outgoing Calls)

ISDN Network-Specific Facilities Information Element

Carrier Identification Code

Network Service Protocol

| Network Service | Service Parameter Name | Service Parameter Value |
|--|---|-------------------------|
| <input type="text" value=" - Not Selected -"/> | <input type="text" value=" < Not Exist >"/> | <input type="text"/> |

Done Local Intranet



Overlap Sending Route Pattern Configuration

Cisco CallManager 3.3 Administration - Route Pattern Configuration - Microsoft Internet Explorer

File Edit View Favorites Tools Help

Back Forward Stop Home Search Favorites History Print

Address <http://ferengi/CCMAdmin/routepatternconfig.asp?plid={D2244682-022B-4947-9D29-4F8714BF86E0}> Go Links

Route Pattern: 9.X

Status: Ready
Note: Any update to this route pattern automatically resets the associated gateway/route list

Copy Update Delete

Pattern Definition

Route Pattern*

Partition

Description

Numbering Plan*

Route Filter

Gateway/Route List* (Edit)

Route Option
 Route this pattern Block this pattern

Provide Outside Dial Tone Urgent Priority

Calling Party Transformations

Use Calling Party's External Phone Number Mask

Calling Party Transform Mask

Prefix Digits (Outgoing Calls)

Calling Party Presentation

Called Party Transformations

Discard Digits

Called Party Transform Mask

Prefix Digits (Outgoing Calls)

ISDN Network-Specific Facilities Information Element

Carrier Identification Code

Network Service Protocol

| Network Service | Service Parameter Name | Service Parameter Value |
|--|---|-------------------------|
| <input type="text" value=" - Not Selected -"/> | <input type="text" value=" < Not Exist >"/> | <input type="text"/> |

Done Local Intranet



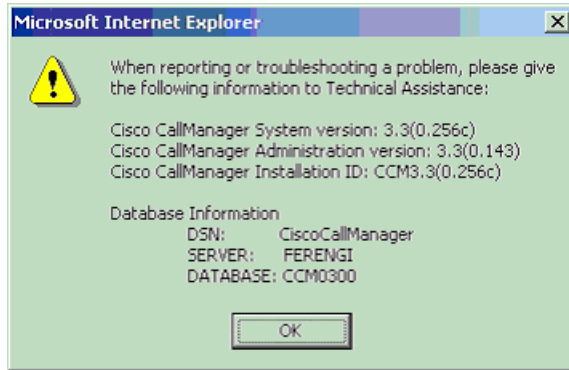
References

Alcatel 4400 Rel4.2 PBX with CCM using 6608-E1 QSIG protocol as MGCP Gateway (Application Note)



Appendix A

CCM Software release:



Alcatel 4400 Software Release

Command: \compidea\node

Table 7 Alcatel Software Parameter Info

| Parameter | Value |
|------------------------------------|----------------|
| Node | |
| Node Number (reserved) | 1 |
| Software Version | R3.2 |
| Version name | c1.712 |
| Patch No. | 5 |
| Notes | |
| Object Identity | |
| Node Number (reserved) | 1 |
| Ethernet Notes\Netmask | |
| Ethernet Notes\Local CPUName | x000000_tun |
| Ethernet Notes\Local CPUIP Address | 172.30.253.253 |
| Ethernet Notes\Twin CpuName | |



Table 7 Alcatel Software Parameter Info

| Parameter | Value |
|---|----------------|
| Ethernet Notes\Twin Cpu\IP Address | |
| Ethernet Notes\Main Cpu\Name | xm000000 |
| Ethernet Notes\Main Cpu\IP Address | 10.253.253.3 |
| Ethernet Notes\StandBy Cpu\Name | |
| Ethernet Notes\StandBy Cpu\IP Address | |
| SL Notes | |
| IP/X25 Tunnel Notes\Netmask | 255.255.0.0 |
| IP/X25 Tunnel Notes\Local Node\Name | x000000_tun |
| IP/X25 Tunnel Notes\Local Node\IP Address | 172.30.253.253 |



Cisco Catalyst 6000 Switch Configuration

```

Console> (enable) sh version
WS-C6506 Software, Version NmpSW: 6.1(4)
Copyright (c) 1995-2001 by Cisco Systems
NMP S/W compiled on May 15 2001, 12:27:20

```

```
System Bootstrap Version: 5.3(1)
```

```
Hardware Version: 2.0 Model: WS-C6506 Serial #: TBA04110341
```

| Mod | Port | Model | Serial # | Versions |
|-----|------|------------------------------|----------------------------|---|
| 1 | 2 | WS-X6K-SUP1A-2GE | SAD041504XL | Hw : 3.1 Fw : 5.3(1) Fw1: 5.1(1)CSX Sw : 6.1(4) Sw1: 6.1(4) |
| 3 | 48 | WS-F6K-PFC WS-X6248-RJ-45 | SAD0413097K SAD04150CK1 | Hw : 1.1 Hw : 1.2 Fw : 5.1(1)CSX Sw : 6.1(4) |
| 5 | 8 | WS-X6608-E1 | SAD043300AJ | Hw : 1.1 Fw : 5.4(2) Sw : 6.1(4) |
| 33) | | | | HP1: D00403030007; DSP1: D005b033 (3.6. |
| 33) | | | | HP2: D00403030007; DSP2: D005b033 (3.6. |
| 33) | | | | HP3: D00403030007; DSP3: D005B033 (3.6. |
| 33) | | | | HP4: D00403030007; DSP4: D005B033 (3.6. |
| 2) | | | | HP5: C00103010007; DSP5: C002E031 (3.3. |
| 2) | | | | HP6: C00103010007; DSP6: C002E031 (3.3. |
| 2) | | | | HP7: C00103010007; DSP7: C002E031 (3.3. |
| 2) | | | | HP8: C00103010007; DSP8: C002E031 (3.3. |

| Module | DRAM | | | FLASH | | | NVRAM | | |
|--------|--------|--------|--------|--------|-------|--------|-------|------|------|
| | Total | Used | Free | Total | Used | Free | Total | Used | Free |
| 1 | 65408K | 43645K | 21763K | 16384K | 5327K | 11057K | 512K | 245K | 267K |

```
Uptime is 199 days, 22 hours, 13 minutes
```

```
Console> (enable)
```

```
Console> (enable) sh module
```

| Mod | Slot | Ports | Module-Type | Model | Sub | Status |
|-----|------|-------|-----------------------|------------------|-----|--------|
| 1 | 1 | 2 | 1000BaseX Supervisor | WS-X6K-SUP1A-2GE | yes | ok |
| 3 | 3 | 48 | 10/100BaseTX Ethernet | WS-X6248-RJ-45 | no | ok |
| 5 | 5 | 8 | E1 | WS-X6608-E1 | no | ok |



```
Mod Module-Name          Serial-Num
-----
1                        SAD041504XL
3                        SAD04150CK1
5                        SAD043300AJ
```

```
Mod MAC-Address(es)      Hw      Fw      Sw
-----
1  00-d0-d3-37-f9-8e to 00-d0-d3-37-f9-8f 3.1     5.3(1)  6.1(4)
   00-d0-d3-37-f9-8c to 00-d0-d3-37-f9-8d
   00-01-63-af-5c-00 to 00-01-63-af-5f-ff
3  00-01-97-4a-10-30 to 00-01-97-4a-10-5f 1.2     5.1(1)CSX 6.1(4)
5  00-01-64-12-22-80 to 00-01-64-12-22-87 1.1     5.4(2)  6.1(4)
```

```
Mod Sub-Type              Sub-Model          Sub-Serial  Sub-Hw
-----
1  L3 Switching Engine    WS-F6K-PFC        SAD0413097K 1.1
```

Console> (enable)

Console> (enable) **sh port 5/1**

```
Port Name                Status      Vlan      Duplex Speed Type
-----
5/1                      connected  1         full  2.048 E1
```

```
Port  DHCP    MAC-Address      IP-Address      Subnet-Mask
-----
5/1   enable  00-01-64-12-22-80 10.10.10.104    255.255.255.0
```

```
Port  Call-Manager(s)  DHCP-Server      TFTP-Server      Gateway
-----
5/1   10.10.10.1       10.10.10.1       10.10.10.1       10.10.10.1
```

```
Port  DNS-Server(s)   Domain
-----
5/1   -               -
```

```
Port  CallManagerState DSP-Type
-----
5/1   registered      C549
```

```
Port  NoiseRegen  NonLinearProcessing
-----
5/1   enabled     enabled
```

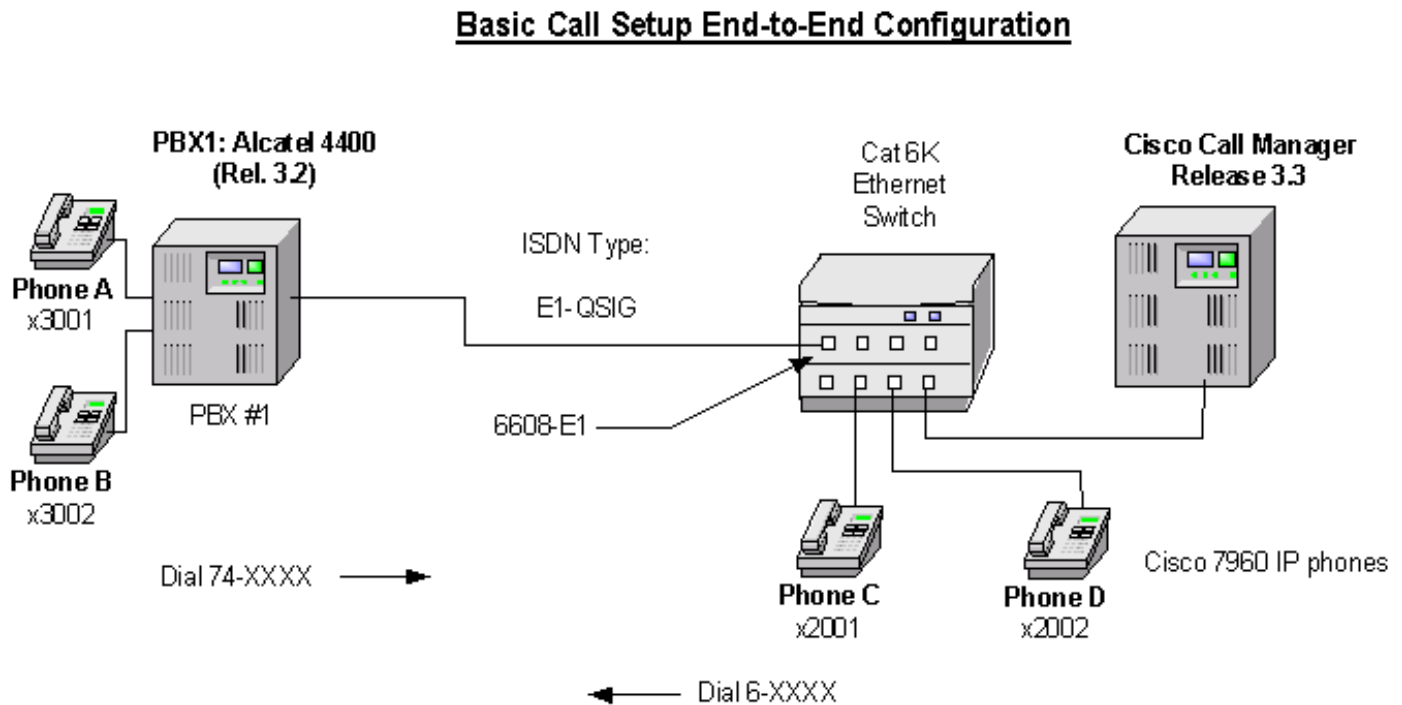
```
Port  Trap      IfIndex
-----
5/1   disabled  66
```

Console> (enable)



Test Configuration

Figure 2
Test Configuration Diagram



As shown in [Figure 2](#), an Alcatel 4400 PBX was connected via an ISDN E1 PRI QSIG link to a Cisco 6608-E1 Gateway, which in turn, was connected to an Ethernet switch. The interoperability testing involved Layers 1, 2 and 3 on the ISDN PRI QSIG link between a Cisco 6608-E1 and the PBX.

Layer 1 (Physical Layer)

The Alcatel 4400 PBX configuration screen for the ISDN PRI QSIG trunk interface is reached using both Alcatel Board and Board/Digital Access menus, setting the PRI physical layer parameters.

Layers 2 & 3 (Q.921 and Q.931)

Layer 2 and 3 packet exchanges were monitored using an Acacia Clarinet protocol analyzer, bridged across the PRI link in high impedance mode.

Layer 2 Q.921 packets were monitored to ensure that each PBX/6608-E1 software configuration properly exchanged SABME/UA packets to initialize the ISDN link, and then RR packets were exchanged every 30 seconds.

Layer 3 Q.931 packets were monitored to ensure that the appropriate call setup/teardown packets were exchanged for each configuration, and that the SETUP packets contained the mandatory Information Elements with the necessary details, as well as optional IEs such as Calling Name and Number.

Telephone calls were made end-to-end in both directions through the Cisco 6608-E1 Gateway, and a check was made to ensure that there was an audio path in both directions for each call.



User/Network Settings

The Cisco 6608-E1 Gateway with ISDN protocol type setting of PRI ISO QSIG E1 supports both protocol sides by selecting “Network/User” in the Protocol Side field when configuring the Gateway via CCM.

The Alcatel 4400, supports both “USER” (slave) and “NETWORK” (master) protocol sides. Network/User options are set in the Board/Digital Access Options menu. Network mode must be set to Yes for (Master/Network) or No - (Slave/User).

The following options for Alcatel PBX are of particular interest:

- Trunk interface type must be set to PRA2.
- Access Type must be set to T2.

Q931 signal variant is used to determine Protocol type. This option was set to ABC-F.



Appendix B

Test Results

Testing was performed by Test Engineer(s): Samir Batio, July 24, 2002

Retesting with CCM3.3-256c load was performed by Test Engineer: Samir Batio, October 1, 2002

Test Setup 1

Test setup was as follows:

- PBX1 configured as ISO version ABC-F, emulates Network.
- Cisco 6608-E1 Gateway configured as PRI ISO QSIG E1, emulates User.

Table 8 Switch and Gateway Settings

| Alcatel 4400 Switch-type / Protocol-Side Setting | Cisco 6608-E1 ISDN protocol-type/ Protocol-Side Setting |
|---|---|
| ABC-F / Network | PRI ISO QSIG E1/User |

Table 9 Basic Calls: (Enbloc Sending)

| Calls Made | Call Comp? | " Calling Number" Displayed on Final Destination? | " Calling Name" Displayed on Final Destination? | " Called Number" Displayed on Orig. Side? | " Called Name" Displayed on Orig. Side? | Notes |
|-----------------------|------------|---|---|---|---|-------|
| Phone A to Phone C | Yes | Yes | Yes | No ¹ | Yes | |
| Phone C to Phone A | Yes | Yes | Yes | Yes | Yes | |

1. Alcatel ISO version ABC-F QSIG link does not honor Connected Number Identification Presentation.



Table 10 Basic Calls with Overlap Sending/Receiving: (check trace to verify Overlap mode)

| Calls Made | Call Comp? | " Calling Number" Displayed on Final Destination? | " Calling Name" Displayed on Final Destination? | " Called Number" Displayed on Orig. Side? | " Called Name" Displayed on Orig. Side? | Notes |
|--------------------|------------|---|---|---|---|-------|
| Phone A to Phone C | Yes | Yes | Yes | No | Yes | |
| Phone C to Phone A | Yes | Yes | Yes | Yes | Yes | 1 |

1. Overlap Receiving on Alcatel PBX works when the Cisco gateway configuration for Called Numbering Plan field is set to 'private', 'unknown', or 'ISDN'.

Table 11 Call Transfers: (Supervised Local Transfers)

| Calls Made | Call Comp? | Orig. " Calling Number" displayed on Final Dest. phone? | Orig. " Calling Name" displayed on Final Dest. phone? | " Called Number" display on Orig. phone updated after transfer? | " Called Name" display on Orig. phone updated after transfer? | Notes |
|-----------------------------------|------------|---|---|---|---|-------|
| Phone C to Phone A Xfr to Phone B | Yes | Yes | Yes | No | No | 1 |
| Phone A to Phone C Xfr to Phone D | Yes | Yes | Yes | No | No | |

1. CM3.3 Seaview release does not support Updating Connected Name and Number for Call Transfers QSIG supplementary Services.



Table 12 Call Transfers: (Supervised Network/External)

| Calls Made | Call Comp? | Orig. "Calling Number" displayed on Final Dest. phone? | Orig. "Calling Name" displayed on Final Dest. phone? | "Called Number" display on Orig. phone updated after transfer? | "Called Name" display on Orig. phone updated after transfer? | Notes |
|-----------------------------------|------------|--|--|--|--|-------|
| Phone C to Phone A Xfr to Phone D | Yes | No | No | No | No | |
| Phone A to Phone C Xfr to Phone B | Yes | No | No | No | No | |

Table 13 Call Conferencing (Local)

| Calls Made | Call Comp? | "Calling Number" displayed on remaining conferee when the conferencing phone drops out? | "Calling Name" displayed on remaining conferee when the conferencing phone drops out? | "Connected Number" updated on Orig. Caller phone display when a conferee drops out? | "Connected Name" updated on Orig. Caller phone display when a conferee drops out? | Notes |
|--|------------|---|---|---|---|--------------|
| Phone C to Phone A, Phone A conf Phone B | Yes | (A Drops out) Yes | (A Drops out) Yes | (A Drops out) No | (A Drops out) No | ¹ |
| Phone C to Phone A, Phone C conf Phone D | Yes | (C Drops out) No | (C Drops out) No | (D Drops out) No | (D Drops out) No | |
| Phone A to Phone C, Phone C conf Phone D | Yes | (C Drops out) No | (C Drops out) No | (C Drops out) No | (C Drops out) No | |
| Phone A to Phone C, Phone A conf Phone B | Yes | (A Drops out) No | (A Drops out) Yes | (B Drops out) No | (B Drops out) Yes | |

1. CM3.3 Seaview release does not honor redirectingName and redirectingNumber IE messages.



Table 14 Call Conferencing (Network/External)

| Calls Made | Call Comp? | " Calling Number" displayed on remaining conferee when the conferencing phone drops out? | " Calling Name" displayed on remaining conferee when the conferencing phone drops out? | " Connected Number" updated on Orig. Caller phone display when a conferee drops out? | " Connected Name" updated on Orig. Caller phone display when a conferee drops out? | Notes |
|--|------------|--|--|--|--|-------|
| Phone C to Phone A, Phone A conf Phone D | Yes | (A Drops out) No | (A Drops out) No | (A Drops out) No | (A Drops out) No | |
| Phone C to Phone A, Phone C conf Phone B | Yes | (C Drops out) Yes | (C Drops out) Yes | (A Drops out) No | (A Drops out) No | |
| Phone A to Phone C, Phone C conf Phone B | Yes | (C Drops out) No | (C Drops out) No | (C Drops out) No | (C Drops out) No | |
| Phone A to Phone C, Phone A conf Phone D | Yes | (A Drops out) No | (A Drops out) No | (C Drops out) No | (C Drops out) Yes | |

Table 15 Call Forward (Local)

| Calls Made | Call Comp? | Original " Calling Number" Displayed on Final Dest.? | Original " Calling Name" Displayed on Final Dest.? | Forwarding " Called Number" Displayed on Final Dest.? | Forwarding " Called Name" Displayed on Final Dest.? | Final dest. " Connec ted Number" updated at orig. side? | Final dest. " Connec ted Name" updated at orig. side? | Notes |
|-----------------------------------|------------|--|--|---|---|---|---|-------|
| Phone C to Phone A fwd to Phone B | No | - | - | - | - | - | - | 1 |
| Phone A to Phone C fwd to Phone D | Yes | Yes | Yes | No | No | No | Yes | |

1. When Alcatel phone is setup for local Call Forwarding it sends Facility IE for 'CallRerouting'. Since CCM3.3 Seaview release does not support 'CallRerouting', the call is not completed.



Table 16 Call Forward (Network/External)

| Calls Made | Call Comp? | Original " Calling Number" Displayed on Final Dest.? | Original " Calling Name" Displayed on Final Dest.? | Forwarding " Called Number" Displayed on Final Dest.? | Forwarding " Called Name" Displayed on Final Dest.? | Final dest. " Connec ted Number" updated at orig. side? | Final dest. " Connec ted Name" updated at orig. side? | Notes |
|-----------------------------------|------------|--|--|---|---|---|---|-------|
| Phone C to Phone A fwd to Phone D | Yes | Yes | No | No | Yes | Yes | No | |
| Phone A to Phone C fwd to Phone B | Yes | Yes | No | No | No | No | Yes | |

Test Setup 2

Setup was as follows:

- PBX1 configured as ISO version ABC-F, emulates User
- Cisco 6608-E1 Gateway configured as PRI ISO QSIG E1, emulates Network

Table 17 Switch and Gateway Settings

| Alcatel 4400 Switch-type/ Protocol-Side Setting | Cisco 6608-E1 ISDN protocol-type/ Protocol-Side Setting |
|---|---|
| ABC-F / User | PRI ISO QSIG E1 / Network |

The test results are the same as in previous section. Refer to Tables 9 through 16 for details.



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